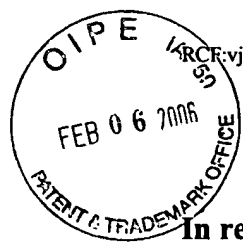


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PATENT

IN THE UNITED STATES PATENT AND TRADEMARK OFFICE

In re application of: Wang et al.

Application No. 10/816,466

Filed: March 31, 2004

Confirmation No. 1042

For: ROBUST REAL-TIME SPEECH CODEC

Examiner: Not yet assigned

Art Unit: 2641

Attorney Reference No. 3382-67640-01

CERTIFICATE OF MAILING

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**INFORMATION DISCLOSURE STATEMENT
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Listed on the accompanying form PTO-1449 and enclosed herewith are several English-language documents. Applicants respectfully request that these documents be listed as references cited on the issued patent.

Copies of United States patents and United States published patent applications do not have to be provided to the Patent Office (37 C.F.R. 1.98(a)(2)(ii)). Copies of unpublished U.S. applications do not have to be provided, as long as the application is available on PAIR, as this requirement of 37 C.F.R. § 1.98(a)(2)(iii) has been waived by the United States Patent and Trademark Office pursuant to the Official Gazette Notice on October 19, 2004 (1287 OG 163). Applicants will provide copies of such patents or applications upon request.

Applicants filed this Information Disclosure Statement ("IDS") before the mailing date of a first Office action on the merits. As a result, no fee should be required to file this IDS.

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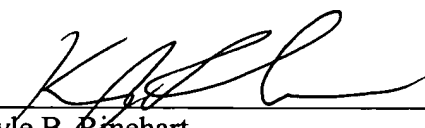
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Respectfully submitted,

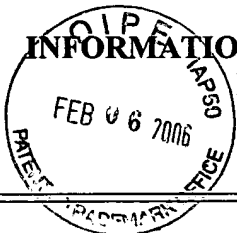
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		WIPO	WO 00 11655	Mar-02	Conexant Systems, Inc.
		WIPO	WO 98 27543	Jun-98	Interval Research Corporation

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